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(56) Documents Cited

EP 0942566 A2

EP 0594358 A2

US 6038261 A

EP 0932285 A2

WO 00/25491 A1

US 5638403 A

(58) Field of Search

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(54) Abstract Title

**Peak power reduction**

(57) In an arrangement for processing a band-limited over-sampled signal (such as a multi-carrier DMT signal, generated by an ADSL modem), the amplitude of portions of the signal having peaks above a threshold are reduced and the frequency distribution of the noise associated with the reduction of such peaks is controlled. As shown a limiter 2 clips a multi-carrier signal to a given threshold, the clipped signals are subtracted 4 from the multi-carrier signal generating clipped samples which are filtered 6 to control the frequency distribution of the clipping noise, and the filtered clipped samples are subtracted 8 from a delayed 10 version of the multi-carrier signal to reduce its peaks.

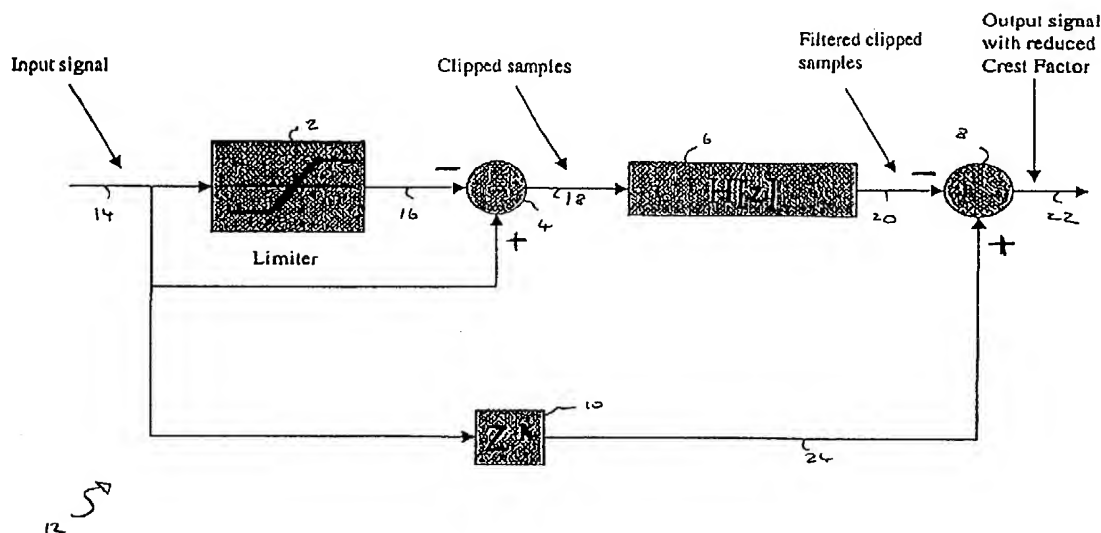


Fig 1

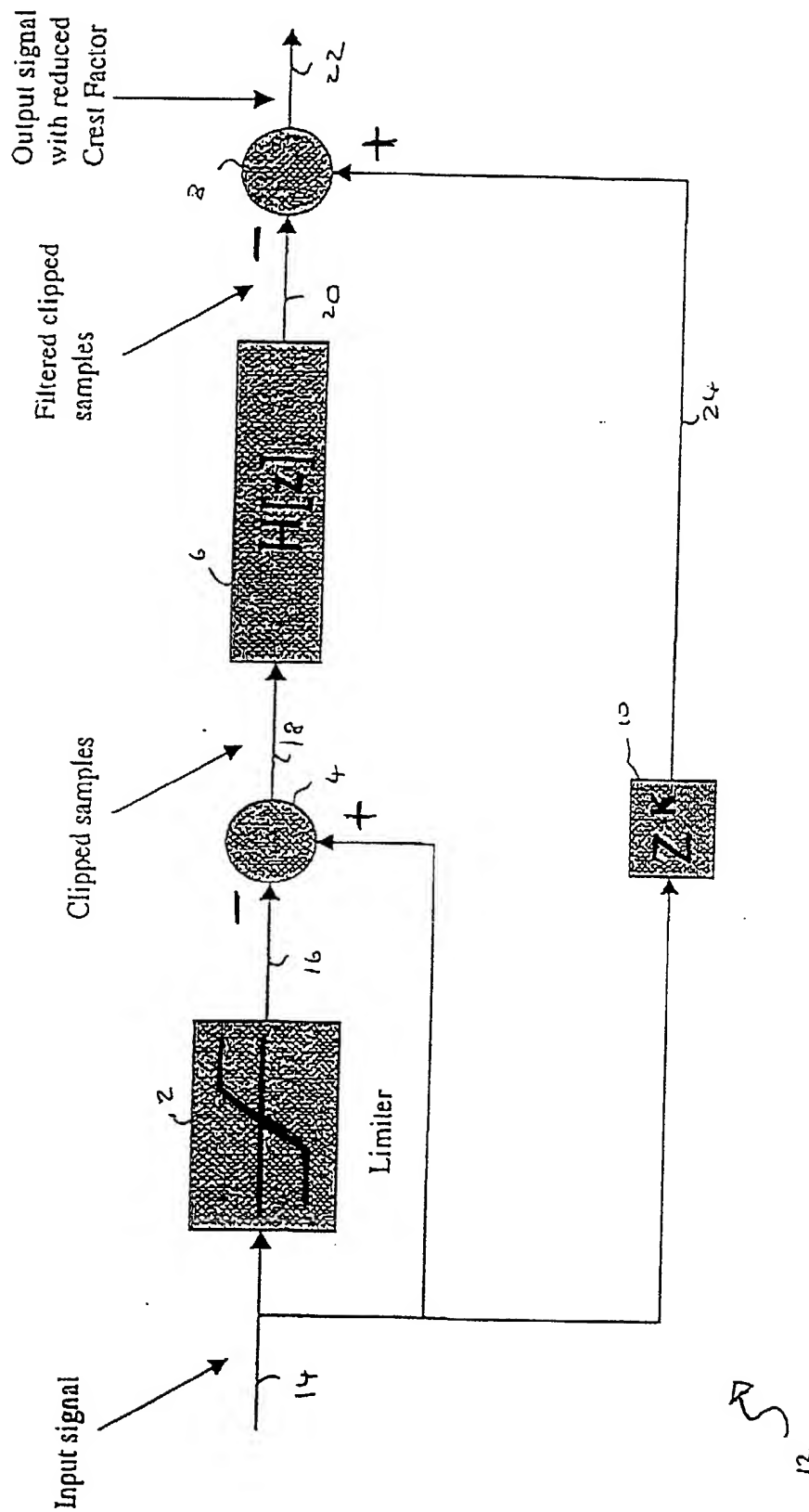
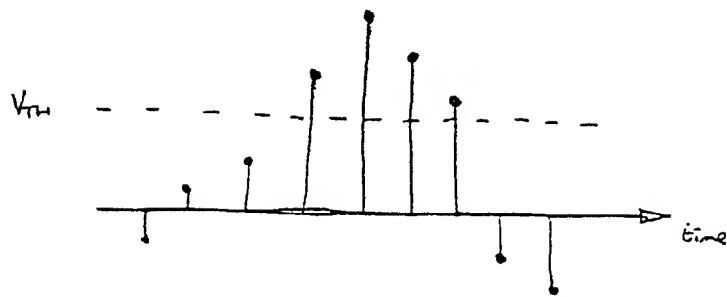
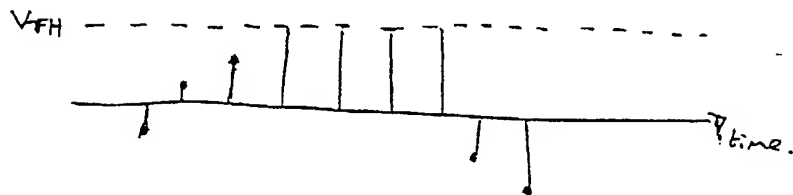
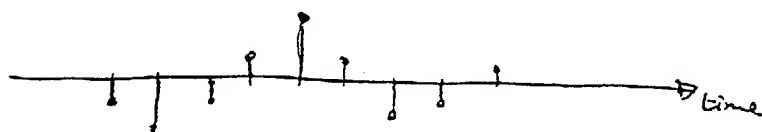
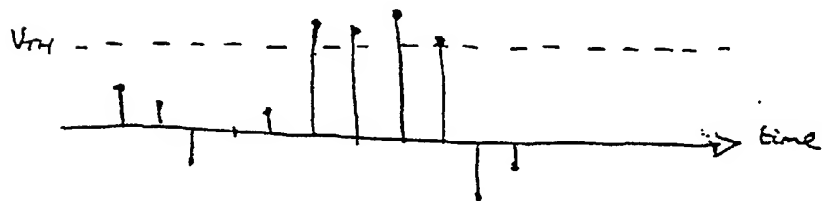


Fig 1

Fig 2.Fig 3Fig 4Fig 5Fig 6

## POWER REDUCTION

The present invention relates to a technique for reducing the maximal signal amplitude and hence the crest factor of a signal, for example, a multi-carrier signal such as a DMT (discrete multi-tone) signal, and particularly but not exclusively to such a signal generated by a modem.

The amplitude distribution of a band-limited, over-sampled signal such as a multi-carrier signal, generated by an ADSL (asymmetric digital subscriber line) modem, can be considered as Gaussian with a zero mean and a variance equal to the power of the transmitter signal. The envelope of such multi-carrier signals randomly present, with a small probability, peaks of a very high amplitude. The span (i.e. the peak to peak voltage) of signals driven by the line driver, which is the component dedicated to transmit the power on the line, is important. The power consumption of the line driver is strongly related to the span of the voltage. In order to keep a reasonable level of power consumption, a certain amount of "clipping" of voltage peaks is desired. Such clipping reduces the voltage peaks of the signal.

However, any such clipping generates broadband noise, usually on the whole bandwidth of the signal. This noise causes loss of data: in the transmit direction by suddenly increasing the background noise; and in the receive direction due to the leakage of the noise through the hybrid. This effect in the receive direction is mainly visible in systems utilising echo cancellers.

Consequently, too frequent clipping causes degradation of the overall bit error ratio of the transmission, resulting in data being lost. Thus, whilst a small amount of clipping is allowable and tolerable, the probability of clipping occurring must be small enough to ensure good performance, i.e. no significant data loss.

US Patent No. 6,038,261 discloses a method for set-up of a  
40 signal in multi-carrier modulation, including clipping of the  
signal amplitude. A feedback loop is utilised to reinject a  
clipping noise from a previous pulse into a new pulse. The  
clipping noise is redistributed outside the useful slip of the  
signal. This method ensures that the signal never exceeds a  
45 predetermined threshold, since the output signal is obtained  
directly from the output of a clipping circuit. However, the  
use of a feedback loop makes the behaviour of the technique  
unpredictable when processing a sequence of pulses. Whilst  
the technique works in a predictable fashion when processing  
50 two successive peaks, feeding the re-distributed noise of the  
first back to the second, with a sequence of peaks operation  
would be unstable. With a sequence of peaks it would be  
difficult to predict the contribution made by successive peaks  
in the feedback loop, and the system would be too  
55 unpredictable for reliable operation.

It is an object of the present invention to provide a  
technique that reduces the maximum signal amplitude of a  
multi-carrier signal whilst maintaining the integrity of the  
60 signal. In particular the invention aims to reduce the  
maximum signal amplitude whilst maintaining the bit error  
ratio of the original multi-carrier signal. Consequently the  
invention reduces the crest factor of the signal.

65 In one aspect the present invention provides a method of  
processing a band-limited, over-sampled signal comprising:  
reducing the amplitude of those portions of the signal having  
peaks above a threshold value; and controlling the frequency  
position of the noise associated with the reduction of such  
70 peaks. Thereby the crest factor of the signal is reduced.  
The amplitude of those portions of the signal having peaks  
above a threshold value are not limited to the threshold  
value. Whilst they may be reduced to a value at or below the  
threshold value, they may be reduced to a level which is above  
75 the threshold value.

The step of reducing the amplitude may comprise: clipping the signal relative to the threshold; filtering the clipped signal; and combining the filtered clipped signal with the  
80 band-limited, over-sampled signal. The step of clipping the signal may comprise: limiting the amplitude of the signal to the threshold; and subtracting the limited signal from the band-limited, over-sampled signal.

85 The step of filtering the clipped signal may comprise creating a pulse having a pre-determined shape dependent upon the clipped samples.

The step of combining may comprise subtracting the filtered  
90 signal from the band-limited, over-sampled signal.

The step of combining may comprise delaying the band-limited, over-sampled signal by an amount corresponding to the time taken to implement the clipping and filtering steps.

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The step of moving the noise may comprise filtering.  
The step of controlling the frequency position of the noise, may comprise moving the noise outside the frequency band used by the signal.

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The step of controlling the frequency position of the noise may comprise moving the noise outside the frequency band used by a signal transmitted in the opposite direction. The signal may contribute an echo to the signal transmitted in the  
105 opposite direction. Although this may result in the clipping noise being present in the used band of the transmitted signal, there may be circumstances where it is more desirable to ensure that no clipping noise contributes to the echo.

110 In another aspect the present invention provides a method of processing a band-limited, over-sampled signal comprising: clipping the signal at a given threshold; subtracting the clipped signal from the signal; filtering the subtracted

115 signal to thereby control the frequency position of the clipping noise; delaying the signal; and subtracting the filtered signal from the delayed signal, thereby reducing the amplitude of those portions of the signal housing peaks above the threshold.

120 In a further aspect the invention provides a circuit for processing a band-limited, over-sampled signal, comprising: circuitry for reducing the amplitude of those portions of the signal having peaks above a threshold value; and circuitry for controlling the frequency position of the noise associated  
125 with such peaks.

The circuitry for reducing may comprise: a limiter for clipping the signal relative to the threshold; a filter for filtering the clipped signal; and an arithmetic unit for  
130 combining the filtered clipped signal with the signal. The arithmetic unit is preferably a subtractor.

The circuit may further include a delay circuit for providing a delayed version of the signal to the arithmetic unit.

135 The circuitry for controlling the frequency position of the noise preferably comprises a filter. The filter preferably controls the frequency position of the noise to move the noise outside the useful frequency band of the signal.

140 The circuitry for controlling the frequency position of the noise may control the noise to be positioned outside the frequency band used by the signal. The circuitry for controlling the frequency position of the noise may control  
145 the noise to be positioned outside the frequency band used by a signal transmitted in the opposite direction. The signal may contribute an echo to the signal transmitted in the opposite direction.

150 In a still further aspect the present invention provides a circuit for processing a band-limited, over-sampled signal

comprising: a limiter for clipping the signal to a given threshold; a first subtractor for subtracting the clipped signal from the multi-carrier signal, thereby generating  
155 clipped samples; a filter for filtering the clipped samples to thereby control the frequency position of the clipping noise; a delay circuit for generating a delayed version of the multi-carrier signal; and a second subtractor for subtracting the filtered clipped samples from the delayed multi-carrier  
160 signal, wherein the amplitude of those portions of the signal having peaks above the threshold is reduced.

The invention will now be described with regard to a non-limiting example with reference to the accompanying drawings  
165 in which:

Figure 1 illustrates a block diagram of a circuit for implementing the technique of the present invention; and

170 Figures 2 to 6 illustrate waveforms at various points in the circuit of Figure 1 in an illustrative example.

The invention is described herein with reference to a particular example concerning the processing of a signal generated by an ADSL (asymmetric digital subscriber line) modem. Such a modem generates a DMT (discrete multi-tone) signal carrying multiple carrier frequencies. Such signals are band-limited, over-sample signals. However, the invention is not limited to the processing of such a signal, and may be  
175 more broadly applied to the processing of any band-limited, over-sampled signal. The invention is not limited in its applicability to multi-carrier signals, but may also apply to single carrier signals. The following description uses the example of a multi-carrier signal for illustrative purposes  
180 only.

As discussed in the introduction hereinabove, as a result of the combining of multiple carriers, the signal at the output of the modem contains random peaks which are above a desired



190 threshold. The level of the desired threshold is user or  
implementation dependent.

Referring to Figure 1, there is illustrated a block diagram of  
a digital circuit suitable for processing the multi-carrier  
195 signal at the output of the modem in accordance with a  
preferred embodiment of the present invention. The thus  
processed signal is then suitable for inputting to a line  
driver.

200 The digital circuit, generally designated by reference numeral  
12, comprises a clipper or limiter 2, a filter 6, a delay unit  
10, and two arithmetic or subtractor units 4 and 8.

The input to the digital circuit 12 on line 14 is, in this  
205 illustrative embodiment, the multi-carrier DMT signal  
comprising samples from the ADSL modem. The samples from the  
ADSL modem are over-sampled in order to obtain a band-limited  
signal. The samples from the ADSL modem on line 14 form an  
input to the limiter 2. Referring to Figure 2, there is  
210 illustrated an example of the samples on line 14 forming an  
input to the digital circuit.

The limiter 2 clips the samples from the ADSL modem on line 14  
at a given threshold and provides clipped samples on line 16.  
215 The threshold applied in the limiter 2 is determined based on  
the tolerable degradation of the bit error ratio, and will be  
user or implementation dependent. In general the lower the  
required bit error ratio, the higher the required threshold.  
The principles and implementation of such clipping are well-  
220 known in the art.

In the present example, it is assumed that the threshold value  
is a level  $V_{TH}$ . The limiter or clipper 2 thus generates a  
sequence of pulses including four pulses corresponding to the  
225 four pulses above the threshold value  $V_{TH}$  as shown in Figure 3.  
As can be seen in Figure 3, the amplitude of these four pulses

is limited at the output of the limiter or clipper 2 to the threshold value  $V_{TH}$ .

230 The clipped samples on line 16 are provided to the arithmetic unit 4 together with the samples from the ADSL modem on line 14. The arithmetic unit 4 subtracts the clipped samples on line 16 from the samples from the ADSL modem on line 14, and a sequence of clipped samples is thus generated on line 18, as  
235 shown in Figure 4.

Referring to Figure 4, it can be seen that the clipped samples indicate four samples corresponding to the four original samples above the threshold value  $V_{TH}$ . The amplitude of these  
240 samples corresponds to their amplitude above the threshold value  $V_{TH}$ .

Thus the subtraction performed by the subtractor 4 extracts the part of the samples of the incoming signals which are  
245 above the threshold.

The sequence of clipped samples on line 18 are then provided as an input to the filter 6. The filter 6 filters the clipped samples on line 18 to shape the noise generated by the  
250 clipping operation. The filter computes a pulse from the clipped samples which has a spectrum outside the signal bandwidth of the useful signal. Thus, in this preferred embodiment, the filter outputs a sequence of samples which have a spectrum outside the used band of the multi-carrier  
255 ADSL signal generated by the modem.

This type of processing, where the samples are moved out of band, is known as soft-clipping. By contrast, hard-clipping would result in the samples (and thus the noise) remaining in  
260 band. Soft-clipping techniques using a filter are well-known to one skilled in the art. Similarly the use of filters generally to control the frequency position of a signal is well-known.

265 In this preferred embodiment the filtering is carried out  
using a Finite Impulse Response Filter.

Although in this preferred embodiment the filter is utilised  
to ensure the noise is out of bound, in general it would be  
270 possible to derive from the clipped samples a pulse with a  
"good" spectrum. However, without shifting the noise outside  
the used frequency band good performance is not achieved in  
the illustrative embodiment described herein.

275 In general terms, the filter controls the frequency position  
of the noise associated with the peaks above the threshold  
value. In certain applications, it may be advantageous to  
control the frequency position within the useful band of the  
band-limited, over-sampled signal.

280 In one example of such a scenario, the noise associated with  
the clipping of the peaks above the threshold is controlled so  
as to be outside the useful band of the signal received in the  
modem. As is well-known in the art, the signal received in the  
285 modem is susceptible to echo effects due to the signal  
transmitted in the same modem, the echo effect being a  
characteristic of a hybrid used to transmit and received  
signals on the communication channel. By ensuring that the  
clipping noise is moved outside the useful band of any signal  
290 received in the hybrid of the modem, then any clipping noise  
leakage in the echo is reduced. Although this may result in  
the clipping noise being present in the used band of the  
transmitted signal, there may be circumstances where it is  
more desirable to ensure that no clipping noise contributes to  
295 the echo.

In the present illustrative embodiment, and in most  
applications, it is preferable to move this noise outside the  
useful band.

300 Thus, in the illustrative embodiment the filter provides a  
modified version of the clipped samples in the incoming signal

in another frequency spectrum. The shape of the filtered clipped samples is shown in Figure 5.

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The sequence of filtered clipped samples generated by the filter 6 are provided on line 20 to the arithmetic unit 8. The arithmetic unit 8 also receives on line 24 a delayed version of the input signal on line 14. The delayed signal is provided by the delay unit 10. The delay is provided to compensate for the computation time in the limiter 2 and the filter 6, such that the timing of the delayed version (or replica) of the input signal coincides with the pulses at the output of the filter. The filtered clipped samples on line 20 are subtracted from the delayed version of the input signal on line 24 in the subtractor 8. This subtraction removes the pulse, generated from the clipped samples, from the incoming signal.

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The waveform resulting from this subtraction is illustrated in Figure 6. As can be seen from Figure 6, although the signal still has peaks above the threshold value  $V_{TH}$ , the peaks of the signal which originally exceeded  $V_{TH}$  are reduced with respect to the original signal. The peak voltage and the power consumption of the signal provided to the line driver on line 22 are thus decreased. Although Figure 6 illustrates that none of the peaks are actually reduced below  $V_{TH}$ , those peaks which are initially only a certain level above  $V_{TH}$  may be reduced to a level at or below  $V_{TH}$ .

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The crest factor of the final signal output by the digital circuit 12 on line 22 depends on three factors:

1. The maximal crest factor of the input signal on line 14;
2. The threshold value used in the limiter 2; and
3. The way in which the pulse is derived from the clipped samples in the filter 6.

335

340

There is an obvious relationship between the first two factors and the final crest factor: the higher the maximal input, the

higher the maximal output; and the higher the threshold value, the higher the maximal output.

345 As regards the filter, optimisation of the design to adjust the final maximal crest factor will be dependent upon the design of the filter itself. The design of the filter is outside the scope of the present invention, but will be readily realised by one skilled in the art.

350 Thus the present invention provides a technique in which any peaks above a threshold level are reduced, but not clipped, such that the effects of such peaks is reduced. Although the implementation of the technique preferably includes a clipping step, it is performed on the front-end rather than as the last  
355 step in the technique, such that the output signal is not a clipped signal. Any noise introduced by the clipping step, so-called clipping noise, is preferably filtered out of the useful frequency band of the signal.

360 Although the invention is described herein with respect to processing samples in the digital domain, the principles of the invention apply also to the analogue domain.

365

CLAIMS:

1. A method of processing a band-limited, over-sampled signal comprising: reducing the amplitude of those portions of  
370 the signal having peaks above a threshold value; and controlling the frequency position of the noise associated with the reduction of such peaks.
2. The method of claim 1 wherein the step of reducing the  
375 amplitude comprises: clipping the signal relative to the threshold; filtering the clipped signal; and combining the filtered clipped signal with the band-limited, over-sampled signal.
- 380 3. The method of claim 2 wherein the step of clipping the signal comprises: limiting the amplitude of the signal to the threshold; and subtracting the limited signal from the band-limited, over-sampled signal.
- 385 4. The method of claim 2 or claim 3 wherein the step of filtering the clipped signal comprises creating a pulse having a pre-determined shape dependent upon the clipped samples.
5. The method of any of claims 2 to 4 wherein the step of combining comprises subtracting the filtered signal from the  
390 band-limited, over-sampled signal.
6. The method of any one of claims 2 to 5 wherein the step of combining comprises delaying the band-limited, over-sampled signal by an amount corresponding to the time taken to  
395 implement the clipping and filtering steps.
7. The method of any one of claims 1 to 6 wherein the step of moving the noise comprises filtering.

8. The method of any preceding claim wherein the step of controlling the frequency position of the noise comprises moving the noise outside the frequency band used by the  
405 signal.

9. The method of any one of claims 1 to 7 wherein the step of controlling the frequency position of the noise comprises moving the noise outside the frequency band used by a signal  
410 transmitted in the opposite direction.

10. The method of claim 9 wherein the signal contributes an echo to the signal transmitted in the opposite direction.

415 11. A method of processing a band-limited, over-sampled signal comprising: clipping the signal at a given threshold; subtracting the clipped signal from the signal; filtering the subtracted signal to thereby control the frequency position of the clipping noise; delaying the signal; and subtracting the  
420 filtered signal from the delayed signal, thereby reducing the amplitude of those portions of the signal having peaks above the threshold.

425 12. A circuit for processing a band-limited, over-sampled signal, comprising: circuitry for reducing the amplitude of those portions of the signal having peaks above a threshold value; and circuitry for controlling the frequency position of the noise associated with the reduction of such peaks.

430 13. The circuit of claim 12 wherein the circuitry for reducing comprises: a limiter for clipping the signal relative to the threshold; a filter for filtering the clipped signal; and an arithmetic unit for combining the filtered clipped signal with the signal.

435 14. The circuit of claim 13 wherein the arithmetic unit is a subtractor.

15. The circuit of any of claims 12 to 14 further including a  
440 delay circuit for providing a delayed version of the signal to  
the arithmetic unit.

16. The circuit of any one of claims 12 to 15 wherein the  
circuitry for controlling the frequency position of the noise  
445 comprises a filter.

17. The circuit of any one of claims 12 to 16 wherein the  
circuitry for controlling the frequency position of the noise  
controls the noise to be positioned outside the frequency band  
450 used by the signal.

18. The circuit of any one of claims 12 to 16 wherein the  
circuitry for controlling the frequency position of the noise  
controls the noise to be positioned outside the frequency band  
455 used by a signal transmitted in the opposite direction.

19. The method of claim 18 wherein the signal contributes an  
echo to the signal transmitted in the opposite direction.

460

20. A circuit for processing a band-limited, over-sampled  
signal comprising: a limiter for clipping the signal to a  
given threshold; a first subtractor for subtracting the  
clipped signal from the multi-carrier signal, thereby  
465 generating clipped samples; a filter for filtering the clipped  
samples to thereby control the frequency position of the  
clipping noise; a delay circuit for generating a delayed  
version of the multi-carrier signal; and a second subtractor  
for subtracting the filtered clipped samples from the delayed  
470 multi-carrier signal, wherein the amplitude of those portions  
of the signal having peaks above the threshold is reduced.





Application No: GB 0017726.1  
Claims searched: 1 to 20

Examiner: Ken Long  
Date of search: 14 February 2001

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## Patents Act 1977 Search Report under Section 17

### Databases searched:

UK Patent Office collections, including GB, EP, WO & US patent specifications, in:

UK Cl (Ed.S): H4P (PAN, PAC, PAR & PAX)  
H4K (KOT & KOD5)

Int Cl (Ed.7): H04L (27/26)

Other: ONLINE : EPODOC, WPI, JAPIO

### Documents considered to be relevant:

Category	Identity of document and relevant passage		Relevant to claims
X	US 6038261	SGS THOMSON (column 2 lines 1-12, column 3 lines 35-42, column 5 lines 5- 9 & 36-39 and column 6 lines 29-43)	1, 2, 7, 8, 12, 13, 16 and 17.
A	EP 0942566 A2	LUCENT	None
A	EP 0932285 A2	TOSHIBA	None
A	EP 0594358 A2	AT&T	None
A	WO 00/25491 A1	PHILIPS N.V.	None
A	US 5638403	MOTOROLA	None

X Document indicating lack of novelty or inventive step  
Y Document indicating lack of inventive step if combined with one or more other documents of same category.

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A Document indicating technological background and/or state of the art.  
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E Patent document published on or after, but with priority date earlier than, the filing date of this application.